



AKSHAYA INSTITUTE OF TECHNOLOGY, TUMKUR
Department of Electronics & Communication Engineering



Module 1 Notes for
“Multimedia Communication”
[BCE613A]

Prepared by:

Mrs. Arshiya Ruheen

Assistant Professor

Department of ECE.

Akshaya Institute of Technology

Tumakuru

AKSHAYA INSTITUTE OF TECHNOLOGY

Lingapura, Obalapura Post, Koratagere Road, Tumakuru - 572106

DEPARTMENT OF ELECTRONICS AND COMMUNICATION ENGINEERING



VISION

To produce competent engineering professionals in the field of Electronics and Communication Engineering by imparting value based quality technical education to meet the societal needs and to develop socially responsible citizens.



MISSION

M1: To provide strong fundamentals and technical skills in the field of Electronics and Communication Engineering through effective teaching learning process.

M2: Enhancing employability of the students by providing skills in the fields of VLSI, Embedded systems, Signal processing, etc., through Centre of Excellence.

M3: Encourage the students to participate in co-curricular and extra-curricular activities that creates a spirit of social responsibility and leadership qualities.



Program Specific Outcomes (PSOs)

After Successful Completion of Electronics and Communication Engineering Program Students will be able to

1. Apply fundamental knowledge of core. Electronics and Communication Engineering in the analysis, design and development of Electronics Systems as well as to interpret and synthesize experimental data leading to valid conclusions.
2. Exhibit the skills gathered to analyze, design, develop software applications and hardware products in the field of embedded systems and allied areas.



Program Educational Objectives (PEOs)

PEO1: Graduates exhibit their innovative ideas and management skills to meet the day to day technical challenges.

PEO2: Graduates utilize their knowledge and skills for the development of optimal solutions to the problems in the field of Electronics and Communication Engineering..

PEO3: Graduates exhibit good interpersonal skills, leadership qualities and adapt themselves for life-long Learning



| | | | |
|---|----------------|------------|-----|
| MultimediaCommunication | | Semester | 6 |
| CourseCode | BCE613A | CIEMarks | 50 |
| TeachingHours/Week(L:T:P:S) | 3:0:0 | SEEMarks | 50 |
| TotalHoursofPedagogy | 40 | TotalMarks | 100 |
| Credits | 03 | ExamHours | |
| Examinationtype(SEE) | Theory | | |
| <p>Courseobjectives:</p> <ul style="list-style-type: none"> ● Gainfundamentalknowledgeinunderstandingthebasicsofdifferentmultimedia Networks and applications. ● Understanddigitizationprincipletechniquesrequiredtoanalyzedifferentmedia Types. ● Analyzecompressiontechniquesrequiredtocompress textandimageandgain Knowledge of DMS. ● Analyzecompressiontechniquesrequiredtocompressaudioandvideo. ● Gainfundamentalknowledgeaboutmultimediaincommunicationacrossdifferent Networks. | | | |
| <p>Teaching-LearningProcess(GeneralInstructions) Theseare sample Strategies, which teachers can use to accelerate the attainment of the various course outcomes.</p> <ol style="list-style-type: none"> 1. Lecture method (L) does not mean only the traditional lecture method, but a different type of teaching method may be adopted to develop the outcomes. 2. Show Video/animation films to explain the functioning of various techniques. 3. Encourage collaborative (Group) Learning in the class. 4. Ask at least three HOTS (Higher-order Thinking) questions in the class, which promote critical thinking 5. Topics will be introduced in multiple representations. 6. Discuss how every concept can be applied to the real world - and when that's possible, it helps improve the students' understanding. | | | |
| Module-1 | | | |
| Multimedia Communications: Introduction , Multimedia information representation, Multimedia networks, multimedia applications, Application and networking terminology. (Chapter 1 of Text 1) | | | |
| Module-2 | | | |
| Information Representation: Introduction, Digitization principles, Text, Images, Audio and Video. (Chapter 2 of Text 1) | | | |
| Module-3 | | | |
| Text and Image Compression: Introduction, Compression principles, text compression, image Compression. (Chapter 3 of Text 1) | | | |
| Module-4 | | | |
| Audio and video compression: Introduction, Audio compression, video compression, video compression principles, video compression. (Chapter 4 of Text 1) | | | |
| Module-5 | | | |
| Multimedia Information Networks: Introduction, LANs, Ethernet, Token ring, Bridges, FDDI (Chapter 8.1 to 8.6 of Text 1). | | | |

Course outcome (Course Skill Set)

At the end of the course, the student will be able to:

1. Understand the basics of multimedia Communication and applications
2. Analyze media type to represent them in digital form.
3. Apply the compression techniques on text, images, audio and video.
4. Understand multimedia information networks.

Assessment Details (both CIE and SEE)

The weightage of Continuous Internal Evaluation (CIE) is 50% and for Semester End Exam (SEE) is 50%. The minimum passing mark for the CIE is 40% of the maximum marks (20 marks out of 50) and for the SEE minimum passing mark is 35% of the maximum marks (18 out of 50 marks). A student shall be deemed to have satisfied the academic requirements and earned the credits allotted to each subject/course if the student secures a minimum of 40% (40 marks out of 100) in the sum total of the CIE (Continuous Internal Evaluation) and SEE (Semester End Examination) taken together.

Continuous Internal Evaluation:

- For the Assignment component of the CIE, there are 25 marks and for the Internal Assessment Test component, there are 25 marks.
- The first test will be administered after 40-50% of the syllabus has been covered, and the second test will be administered after 85-90% of the syllabus has been covered
- Any two assignment methods mentioned in the 2.2 OB 2.4, if an assignment is project-based then only one assignment for the course shall be planned. The teachers should not conduct two assignments at the end of the semester if two assignments are planned.
- For the course, CIE marks will be based on a scaled-down sum of two tests and other methods of assessment.

Internal Assessment Test question paper is designed to attain the different levels of Bloom's taxonomy as per the outcome defined for the course.

Semester-End Examination:

Theory SEE will be conducted by University as per the scheduled timetable, with common question papers for the course **(duration 03 hours).**

1. The question paper will have ten questions. Each question is set for 20 marks.
2. There will be 2 questions from each module. Each of the two questions under a module (with a maximum of 3 sub-questions), **should have a mix of topics** under that module.
3. The students have to answer 5 full questions, selecting one full question from each module.
4. Marks scored shall be proportionally reduced to 50 marks

Suggested Learning Resources:

Textbooks:

Multimedia Communications – Fred Halsall, Pearson Education, 2001, ISBN-978813170994

Reference Books:

1. Multimedia: Computing, Communications and Applications - Raif Steinmetz, Klara Nahrstedt, Pearson Education, 2002, ISBN-978817758
2. Fundamentals of Multimedia - Ze-Nian Li, Mark S Drew, and Jiangchuan Liu.

Web links and Video Lectures (e-Resources):

- Implementation of compression algorithms using MATLAB/any open source tools (Python, Scilab, etc.)

ActivityBasedLearning(SuggestedActivitiesinClass)/PracticalBasedlearning

- <https://www.slideshare.net>
NPTELVideoLectures
- <https://archive.nptel.ac.in/courses/117/105/117105083/>
- MultimediaComputinglecture:Communications&Networking–YouTube

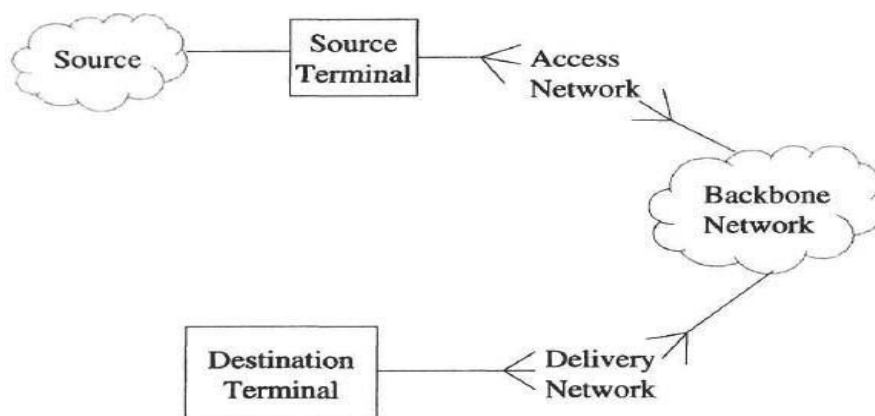
Module1: Multimedia Communications

Multimedia communications have emerged as a major research and development area. In particular, computers in multimedia open a wide range of possibilities by combining different types of digital media such as text, graphics, audio, and video. The emergence of the World Wide Web (WWW), two decades ago, has fuelled the growth of multimedia computing.

Multimedia – an interactive presentation of speech, audio, video, graphics, and text, has become a major theme in today's information technology that merges the practices of communications, computing, and information processing into an interdisciplinary field. In recent years, there has been a tremendous amount of activity in the area of multimedia communications: applications, middleware, and networking. A variety of techniques from various disciplines such as image and video processing, computer vision, audio and speech processing, statistical pattern recognition, learning theory, and data-based research have been employed.

In this chapter, we are interested in multimedia communications; that is, we are interested in the transmission of multimedia information over networks. By *multimedia*, we mean data, voice, graphics, still images, audio, and video, and we require that the networks support the transmission of multiple media, often at the same time.

Fig1.1: components of multimedia communication network



In Figure 1.1 the Source consists of any one or more of the multimedia sources, and the job of the Source Terminal is to compress the Source such that the bit rate delivered to the network connection between the Source Terminal and the Destination Terminal is at least approximately appropriate. Other factors may be considered by the Source Terminal as well. For example, the Source Terminal may be a battery-power-limited device or may be aware that the Destination Terminal is limited in signal processing power or display capability.

Further, the Source Terminal may packetize the data in a special way to guard against packet loss and aid error concealment at the Destination Terminal. All such factors impinge on the design of the Source Terminal. The Access Network may be reasonably modeled by a single line connection, such as a 28.8 Kbit/s modem, a 56 Kbit/s modem, a 1.5 Mbit/s Asymmetric Digital Subscriber Line (ADSL) line, and so on, or it may actually be a network that has shared capacity, and hence have packet loss and delay characteristics in addition to certain rate constraints. The Backbone Network may consist of a physical circuit-switched connection, a dedicated virtual path through a packet-switched network, or a standard best-effort Transmission Control Protocol/Internet Protocol (TCP/IP) connection, among other possibilities. Thus, this network has characteristics such as bandwidth, latency, jitter, and packet loss, and may or may not have the possibility of Quality of Service (QoS) guarantees. The Delivery Network may have the same general set of characteristics as the Access Network, or one may envision that in a one-to-many transmission that the Delivery Network might be a corporate intranet.

Finally, the Destination Terminal may have varying power, mobility, display or audio capabilities.

- —Multimedia indicate that the information/data being transferred over the network may be composed of one or more of the following media types:
 - Text
 - Images
 - Audio
 - video

- Mediatypes
 - Text:unformattedtext,formattedtext
 - Images:computer-generated,pictures
 - Audio:speech,music,generalaudio
 - Video:videoclips,movies,films
- Networktypes
- Multimedia + Network →multimediacommunications

Multimedia Information Representation

- Text,images
 - Blocksofdigitaldata
 - Doesnotvarywithtime(time-independent)
 - Audio,video
 - Varywithtime(time-dependent)
 - Analogsignal
 - Mustbeconvertedintodigitalformforintegration
- Communicationnetworkscannotsupportthehighbitratesofaudio,video
→Compressionisappliedtodigitizedsignals.

MultimediaNetworks:

Many applications, such as video mail, video conferencing, and collaborative work systems,requirenetworkedmultimedia.Intheseapplications,themultimedia objects are stored at a server and played back at the client's sites. Such applications mightrequirebroadcastingmultimediatodatovariousremotelocationsoraccessing large depositories of multimedia sources. Multimedia networks require a very high transferrateorbandwidth,evenwhentheataiscompressed.Traditionalnetworks areusedtoprovideerror-free transmission.However,mostmultimediaapplications can tolerate errors in transmission due to corruption or packet loss without retransmissionorcorrection.Insomecases,tomeetreal-timedeliveryrequirements ortoachievesynchronization,somepackets areevendiscarded.As a result, we can

apply lightweight transmission protocols to multimedia networks. These protocols cannot accept retransmission, since that might introduce unacceptable delays.

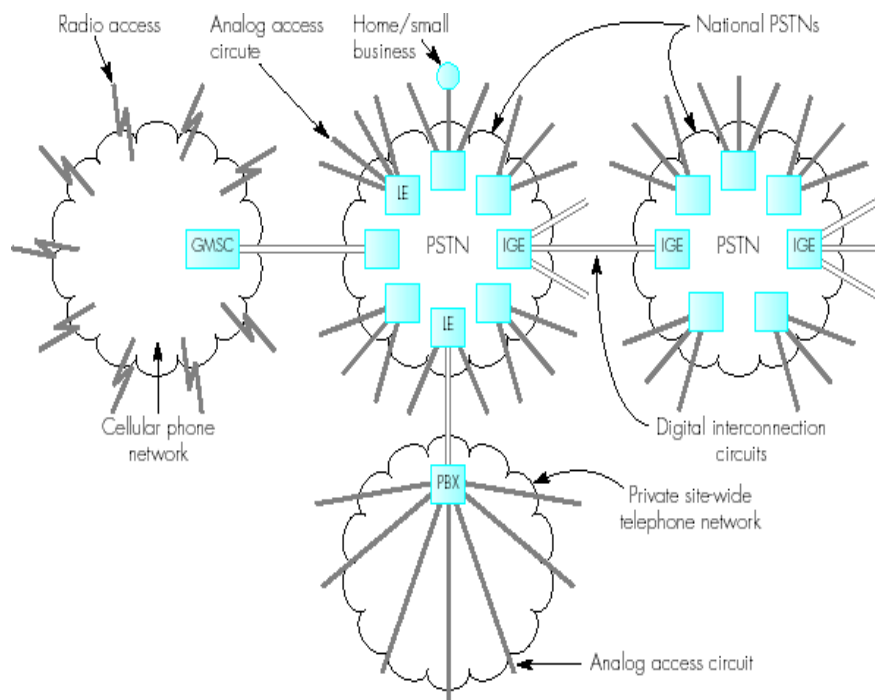
Multimedia networks must provide the low latency required for interactive operation. Since multimedia data must be synchronized when it arrives at the destination site, networks should provide synchronized transmission with low jitter. In multimedia networks, most communications are multipoint as opposed to traditional point-to-point communication. For example, conferences involving more than two participants need to distribute information in different media to each participant.

Conference networks use multicasting and bridging distribution methods. Multicasting replicates a single input signal and delivers it to multiple destinations. Bridging combines multiple input signals into one or more output signals, which then deliver to the participants.

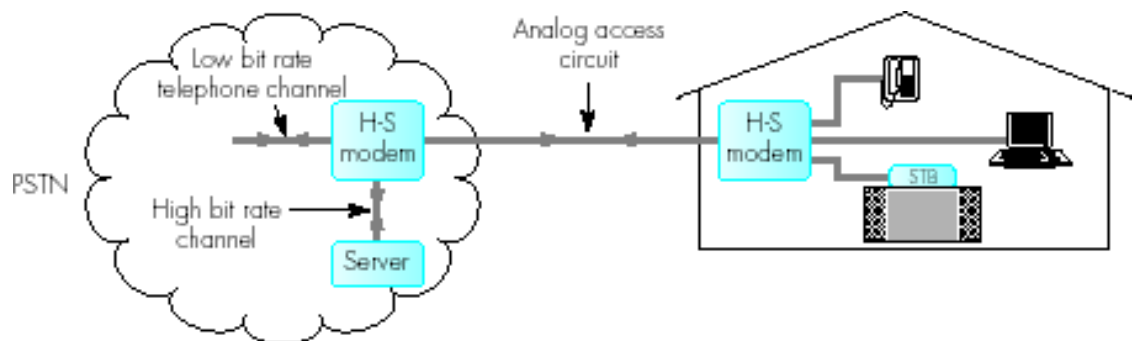
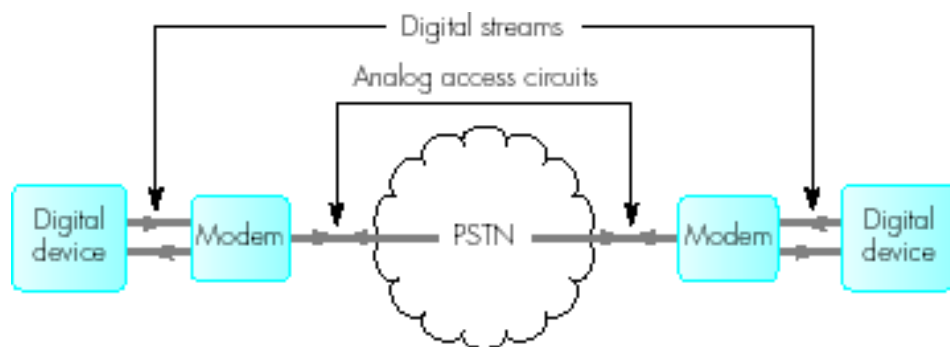
Traditional networks do not suit multimedia Ethernet, which provides only 10 Mbps, its access time is not bounded, and its latency and jitter are unpredictable. Token-ring networks provide 16 Mbps and are deterministic. From this point of view, they can handle multimedia. However, the predictable worst case access latency can be very high.

A fiber distributed data interface (FDDI) network provides 100 Mb/s bandwidth, sufficient for multimedia. In the synchronized mode, FDDI has a low access latency and low jitter. It also guarantees a bounded access delay and a predictable average bandwidth for synchronous traffic. However, due to the high cost, FDDI networks are used primarily for backbone networks, rather than networks of workstations.

• Telephonetworks



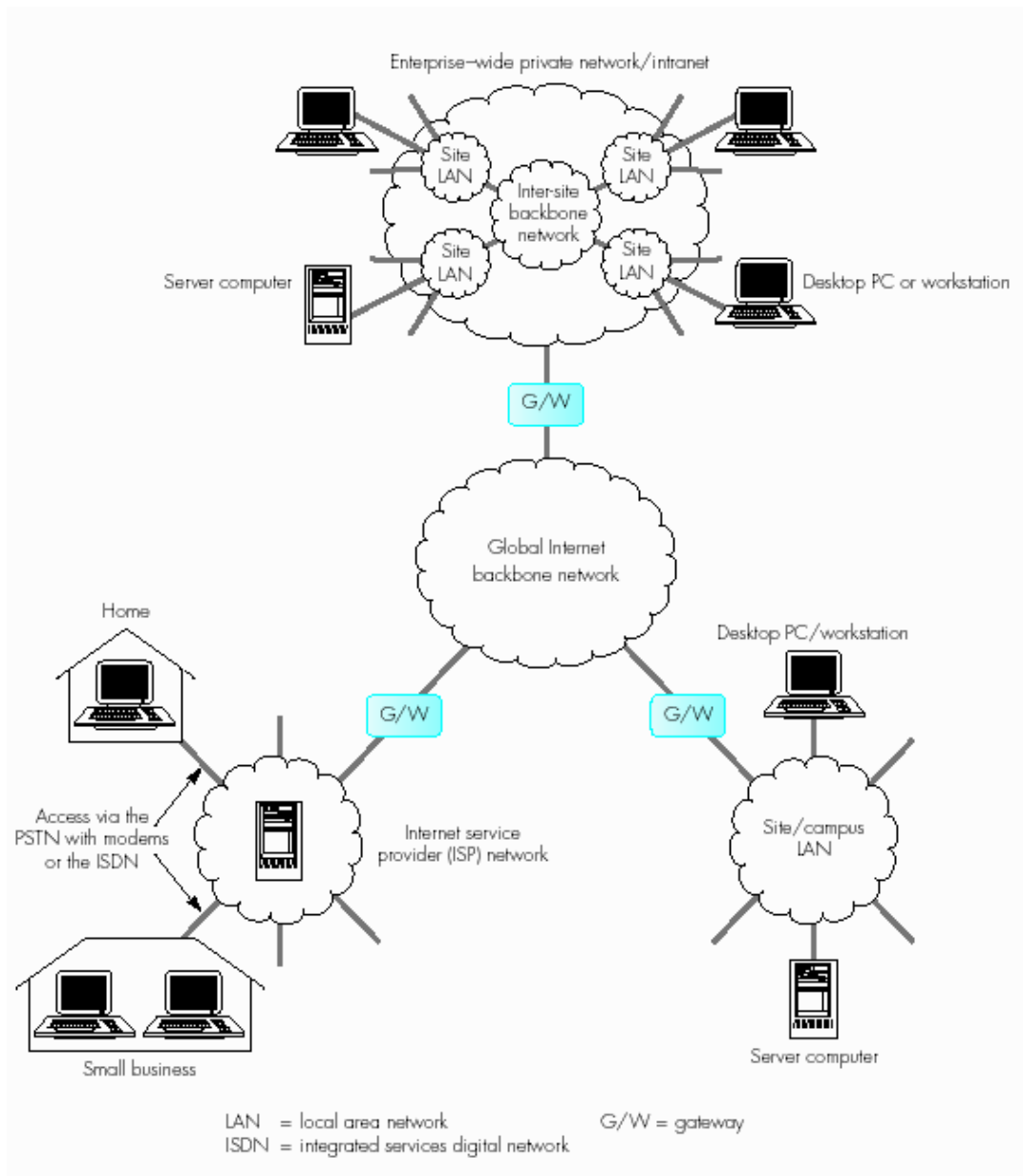
PSTN = public switched telephone network
 GMSC = gateway mobile switching center
 IGE = international gateway exchange
 IE = local exchange/end office
 PBX = private branch exchange



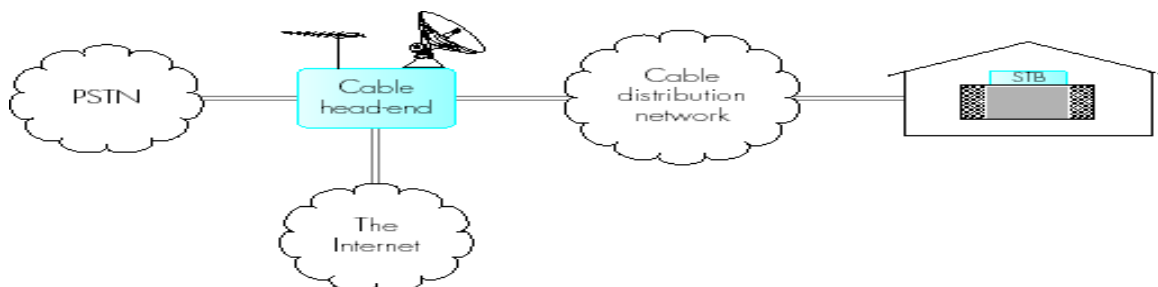
H-S = high-speed

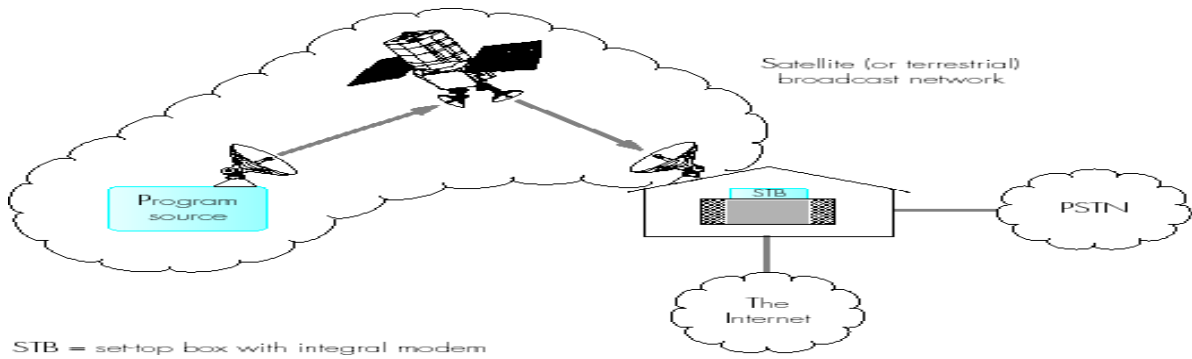
STB = settop box

- **Datanetworks:**

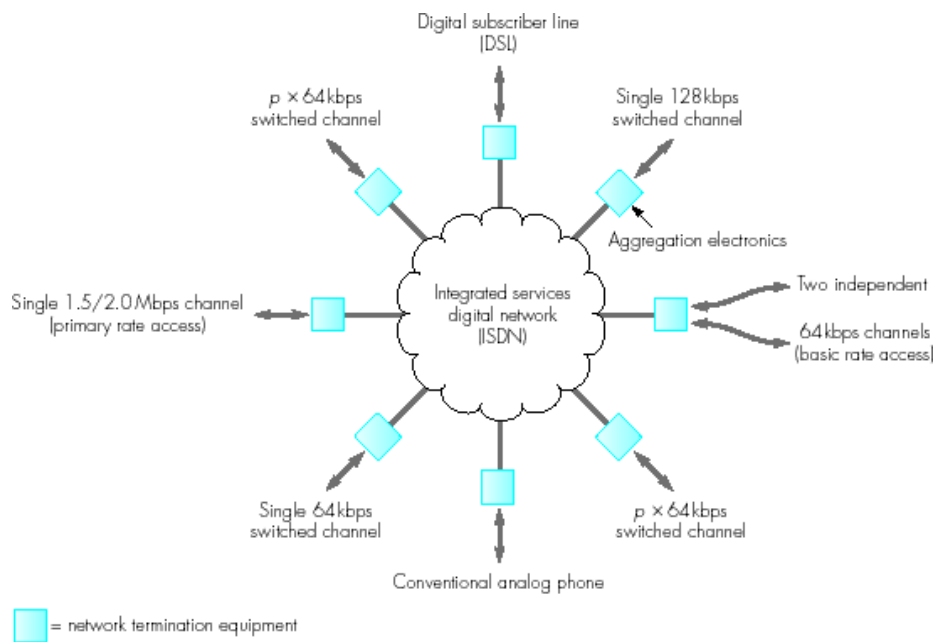


- **Broadcasttelevisionnetworks**

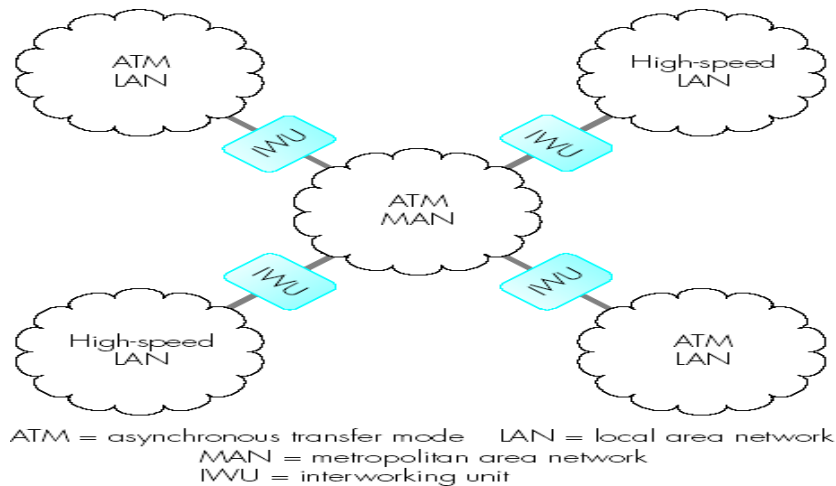




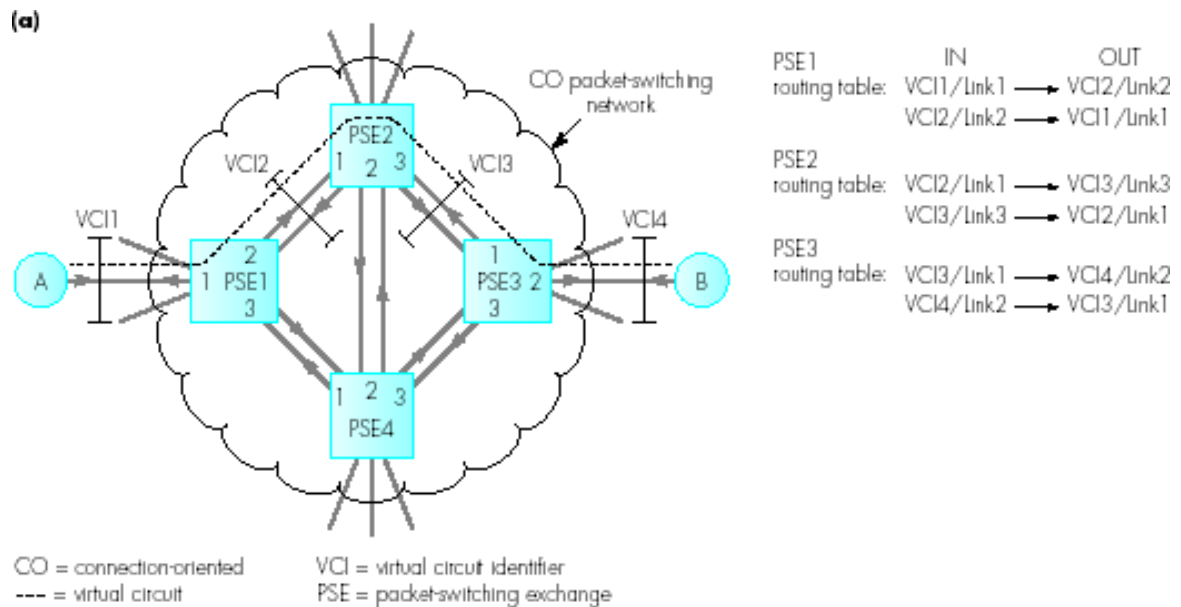
▪ Integrated services digital networks



▪ Broadcast multiservice networks

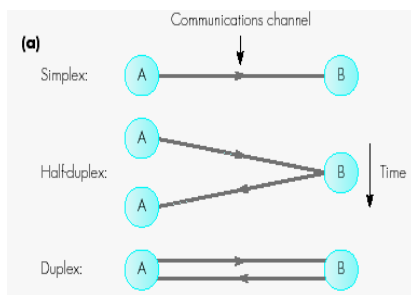


CO packet-switched network including routing table:



Communication modes:

(a) unicast



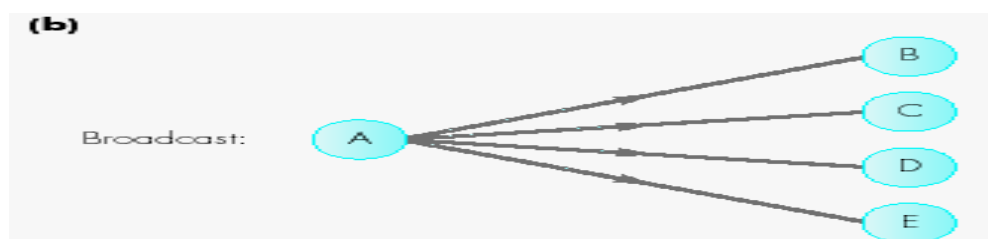
▪ Simplex:

Information flows in one direction only
 Ex) transmission of images from deep space probe

▪ Half-duplex: two-way alternate information flows in both directions but alternatively
 Ex) remote server

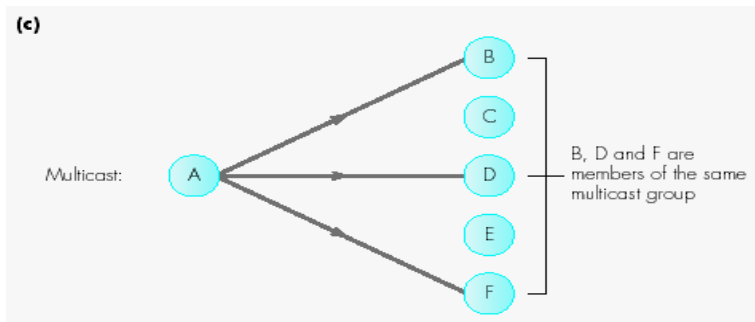
▪ Duplex: two-way simultaneous information flows in both directions simultaneously
 Ex) video telephony

(b) Broadcast:



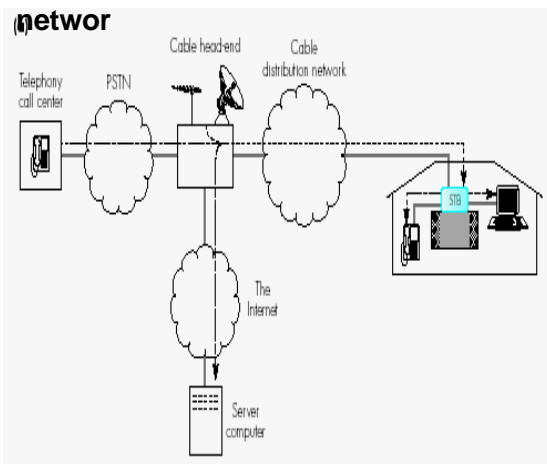
Information output by a single source is received by all other nodes. Ex) cable program over cable network.

c) Multicast: Information output by source is received by specific nodes multicast group Ex) video conferencing

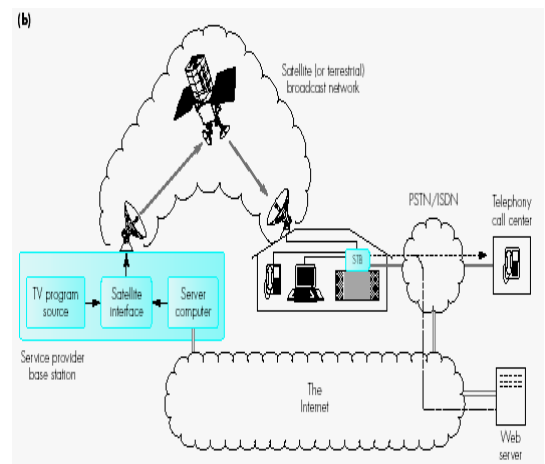


Interactivetelevision:

(a) **cabledistributionnetwork**



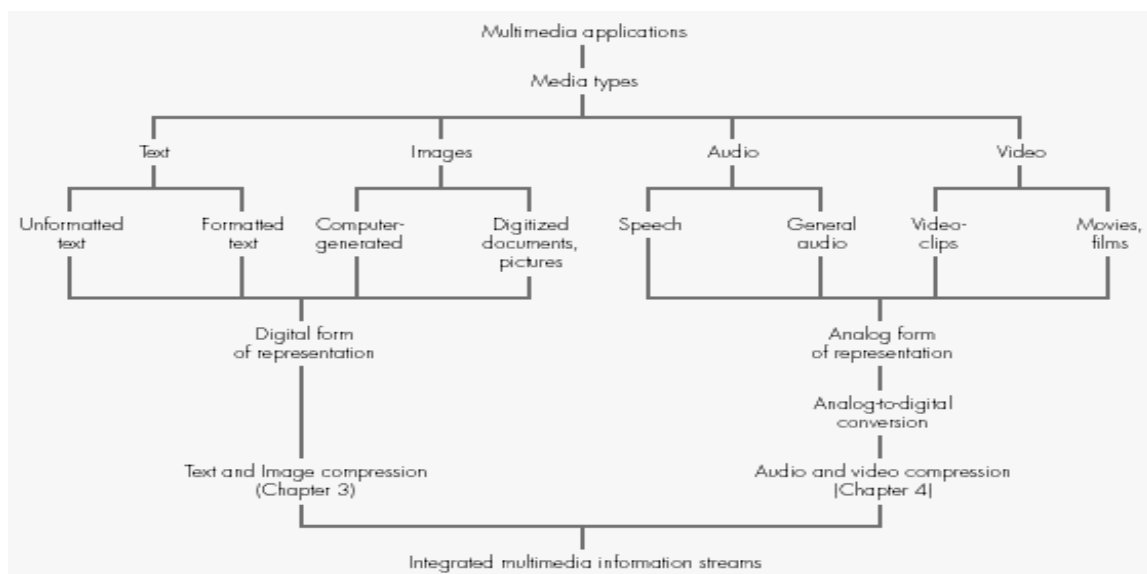
(b) **satellite/terrestrial broadcast**



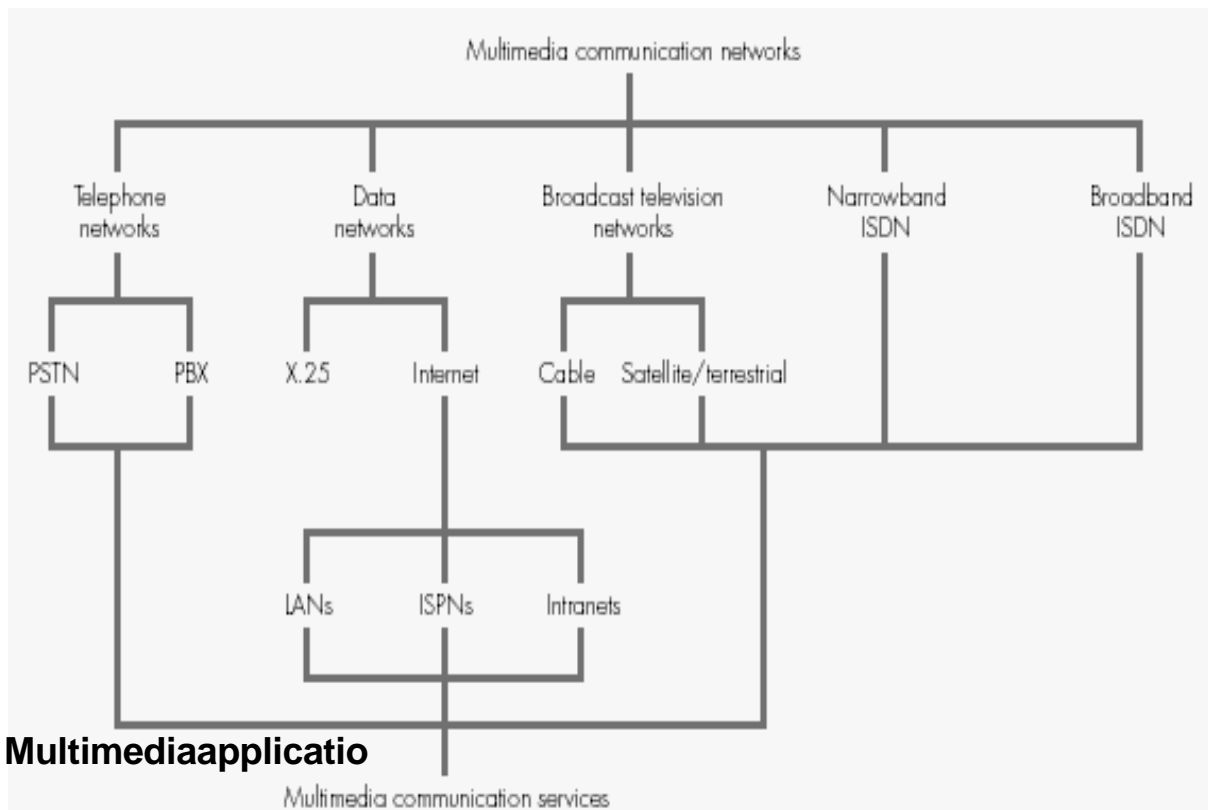
Cablenetwork

- STB provides both low bit rate connection to PSTN and high bit rate connection to Internet
- Subscriber is able to actively respond to the information being broadcast through PSTN
- Typical return channel use: voting, game, homeshopping
- Satellite and terrestrial broadcast networks
- STB provides similar service

MediaTypes



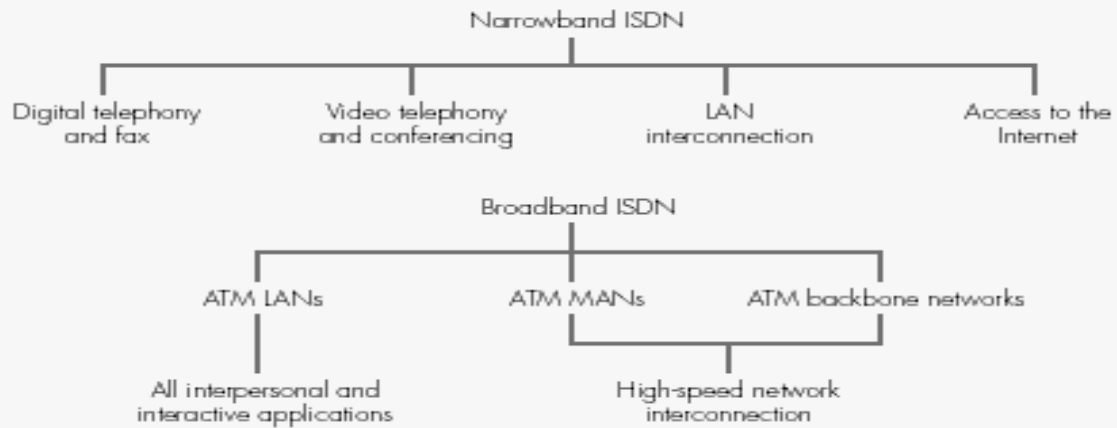
NetworkTypes



Multimedia applicatio

PSTN = public switched telephone network
 PBX = private branch exchange
 ISDN = integrated services digital network

LANs = local area networks
 ISPs = internet service provider networks



Recommended questions:

1. Define Multimedia.?[02]
2. List some of Multimedia applications.[05]
3. List the multimedia communication networks.[05]
4. State Nyquist sampling theorem & Nyquist rate.[05]
5. What do you mean by Hypertext?[04]
6. What is compression?[03]
7. Compare lossy & lossless compression.[07]
8. State & explain the basic form of representation of: Text, Image, Audio, Video?[10]
9. Explain the meaning of bps in relation to digitized audio & video.[05]
10. Explain the meaning of compression & why it is used?[08]
11. Explain the meaning of POTS, local exchange office, PBX, mobile switching centre, international gateway exchange.[08]
12. Explain why most data networks operate in a packet mode. Hence explain why services involving audio and video are supported?[10]
13. Explain — Broadband in relation to B-ISDN and why deployment has been delayed?[08]
14. Describe the principal operation of a fax machine & why modems are required. What is the meaning of PC fax?[10]
15. With aid of block diagram explain CSCW?[10]
16. Explain a Web server, a browser, WWW?[05]
17. Explain webpage, homepage, hyperlink, URL, HTML?[10]

Module 2: Information Representation

Introduction

- The conversion of an analog signal into a digital form
- Signal encoder, sampling, signal decoder

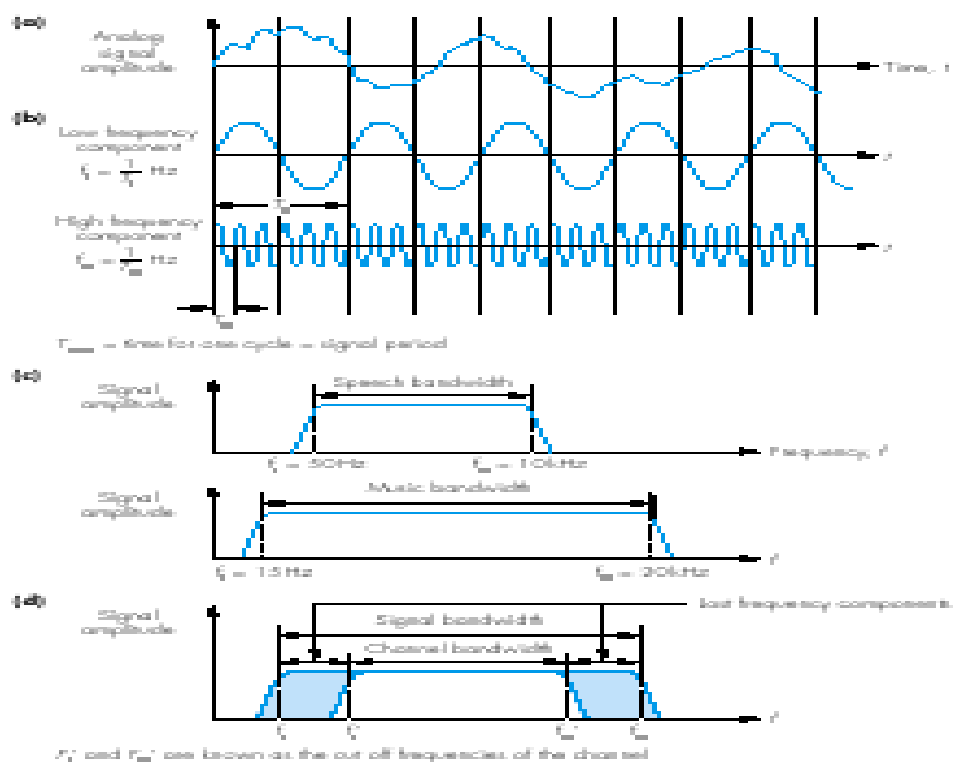
Digitization

principles

Analog signals

- Fourier analysis can be used to show that any time-varying analog signal is made up of a possibly infinite number of single-frequency sinusoidal signals whose amplitude and phase vary continuously with time relative to each other
- Signal bandwidth
- Fig 2.1
- The bandwidth of the transmission channel should be equal to or greater than the bandwidth of the signal—band limiting channel

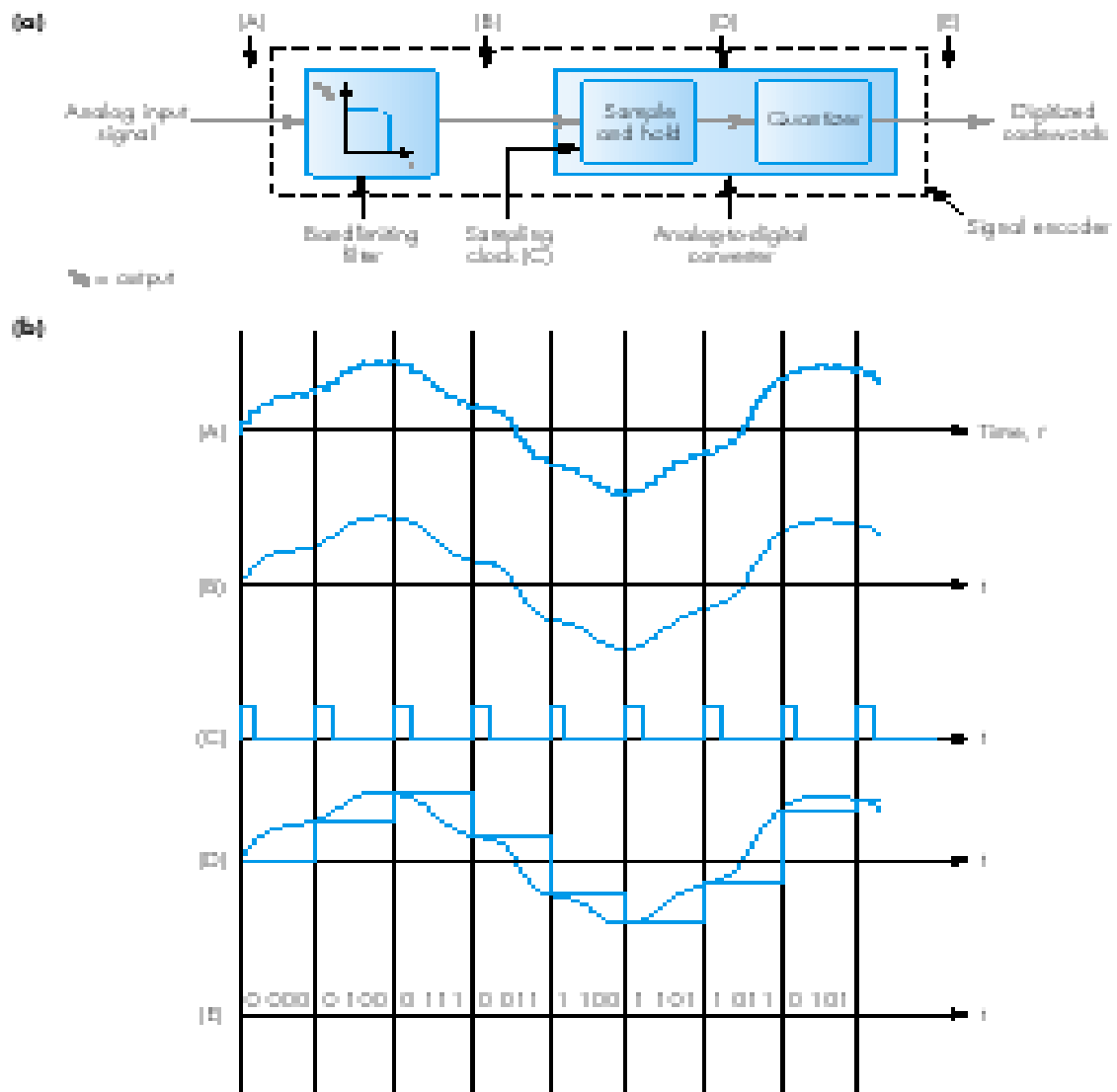
Figure 2.1 Signal properties: (a) time-varying analog signal; (b) sinusoidal frequency components; (c) signal bandwidth examples; (d) effect of a limited bandwidth transmission channel.



Encoder design

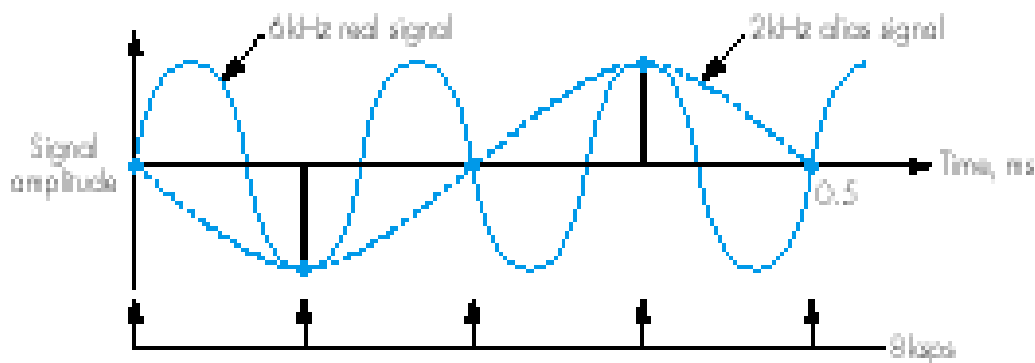
- A bandlimiting filter and an analog-to-digital converter (ADC), the latter comprising a sample-and-hold and a quantizer
- Fig 2.2
- Remove selected higher-frequency components from the source signal (A)
- (B) is then fed to the sample-and-hold circuit
- Sample the amplitude of the filtered signal at regular time intervals (C) and hold the sample amplitude constant between samples (D)

Figure 2.2 Signal encoder design: (a) circuit components; (b) associated waveform set.



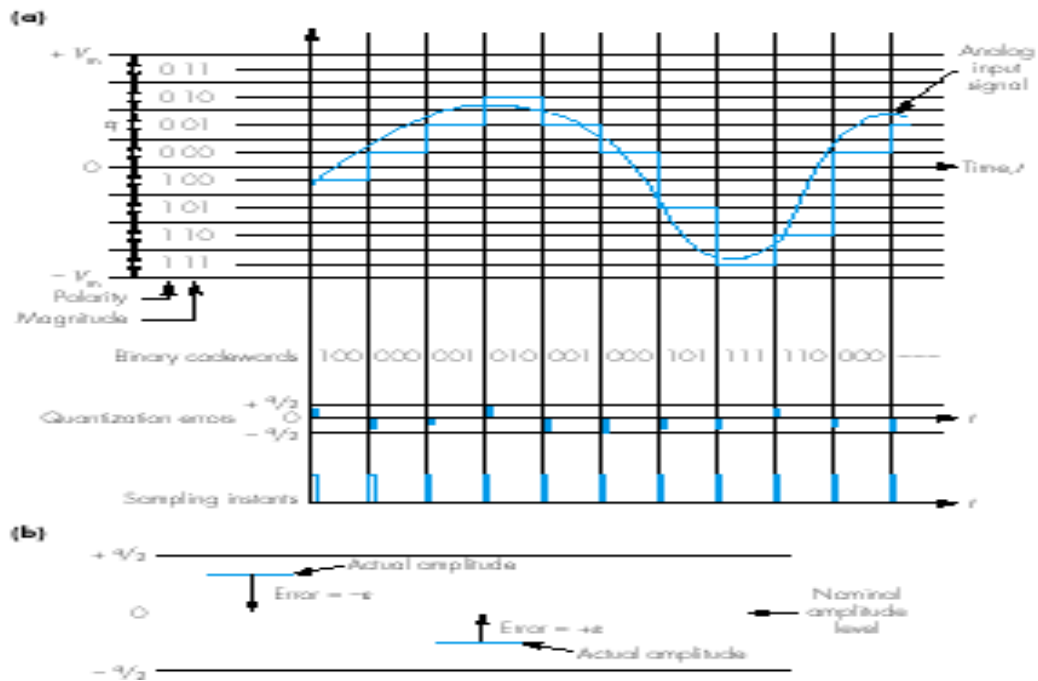
- Quantizer circuit which converts each sample amplitude into a binary value known as a codeword (E)
- The signal to be sampled at a rate which is higher than the maximum rate of change of the signal amplitude
- The number of different quantization levels used to be as large as possible
- Nyquist sampling theorem states that: in order to obtain an accurate representation of a time-varying analog signal, its amplitude must be sampled at a minimum rate that is equal to or greater than twice the highest sinusoidal frequency component that is present in the signal
- Nyquist rate: samples per second (sps)
- The distortion caused by sampling a signal at a rate lower than the Nyquist rate
- Fig 2.3
- Aliassignals: they replace the corresponding original signals

Figure 2.3 Aliassignal generation due to undersampling.



- Quantization intervals
- A finite number of digits is used, each sample can only be represented by a corresponding number of discrete levels
- Fig 2.4
- If V_{max} is the maximum positive and negative signal amplitude and n is the number of binary bits used, then the magnitude of each quantization interval, q

Figure 2.4 Quantization procedure: (a) source of errors; (b) noise polarity.



- Each codeword corresponds to a nominal amplitude level which is at the center of the corresponding quantization interval
- The difference between the actual signal amplitude and the corresponding nominal amplitude is called the quantization error (Quantization noise)
- The ratio of the peak amplitude of a signal to its minimum amplitude is known as the dynamic range of the signal, D (decibels or dB)

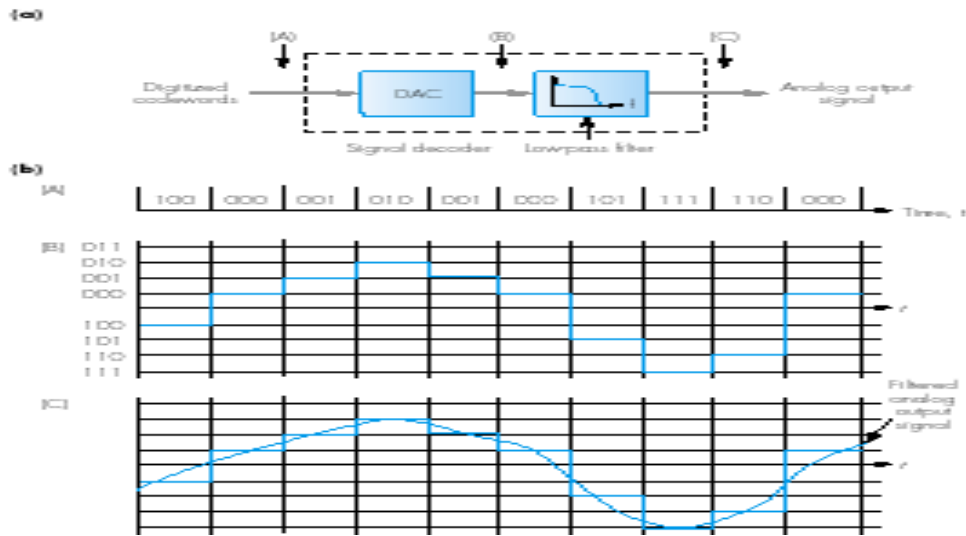
$$D = 20 \log \left(\frac{V_{\max}}{V_{\min}} \right) \text{ dB}$$

- It is necessary to ensure that the level of quantization noise relative to the smallest signal amplitude is acceptable
- Example 2.2

Decoder design

- Fig 2.5
- Reproduce the original signal, the output of the DAC is passed through a low-pass filter which only passes those frequency components that made up the original filtered signal (C)
- Audio/video encoder-decoder or audio/video codec

Figure 2.5 Signal decoder design: (a) circuit components; (b) associated waveform set.



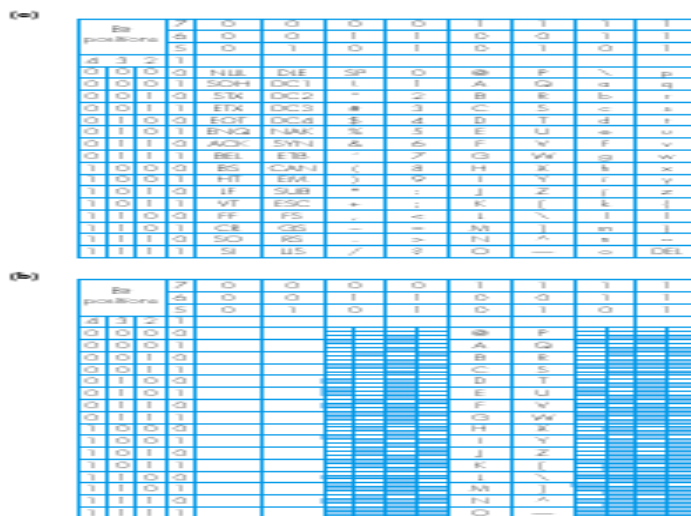
Text

- Threetypesof text
 - Unformatted text
 - Formatted text
 - hypertext

Unformatted text

- American Standard Code for Information Interchange (ASCII character set)
- Fig 2.6
- Mosaic characters create relatively simple graphical images

Figure 2.6 Two example character sets to produce unformatted text: (a) the basic ASCII character set; (b) supplementary set of mosaic characters.



Formatted text

- Produced by most word processing packages
- Each with different headings and with tables, graphics, and pictures inserted at appropriate points
- Fig 2.8
- WYSIWYG: an acronym for what-you-see-is-what-you-get

Figure 2.8 Formatted text: (a) an example of formatted text string; (b) printed version of the string.

(a)

```
<B><FONT SIZE=4><P>Formatted Text</P>
</B></FONT>
<P>This is an example of formatted text. It includes:
<FONT SIZE=2>
</FONT><I><P>italics,</I> <B>bold</B> and <U>underlining</U>
</P>
<FONT FACE="French Script MT"><P>Different Fonts</P></FONT> and <FONT
SIZE=4>Font Sizes</P>
```

(b)

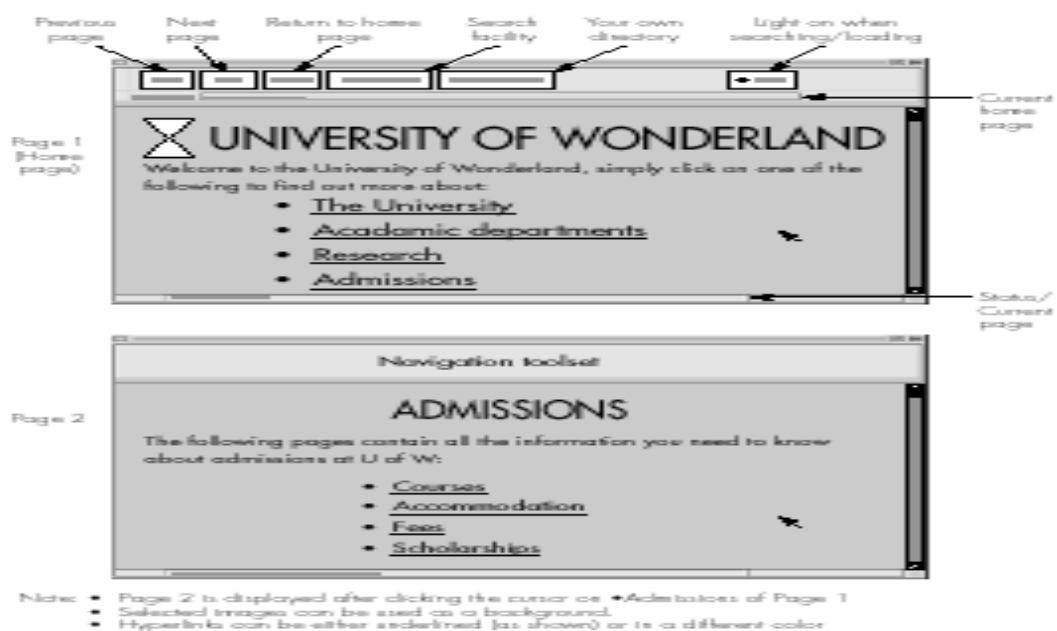
Formatted text

This is an example of formatted text. It includes:
 Italics, **Bold** and Underlining
 Different fonts and Font Sizes

Hypertext

- Formatted text that enables a related set of documents—normally referred to as pages—to be created which have defined linkage points—referred to as hyperlinks—between each other
- Fig 2.9

Figure 2.9 Example of an electronic document written in hypertext.



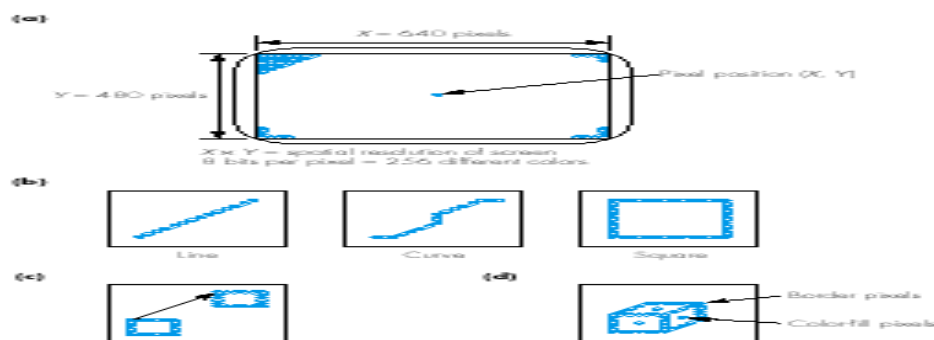
Images

- Image are displayed in the form of a two-dimensional matrix of individual picture elements—known as pixels or pels

Graphics

- Fig2.10
- Two forms of representation of a computer graphic: a high-level version (similar to the source code of a high-level program) and the actual pixel- image of the graphic (similar to the byte-string corresponding to the low-level machine code—bit-map format)
- Standardized forms of representation such as GIF (graphical interchange format) and TIFF (tagged image file format)

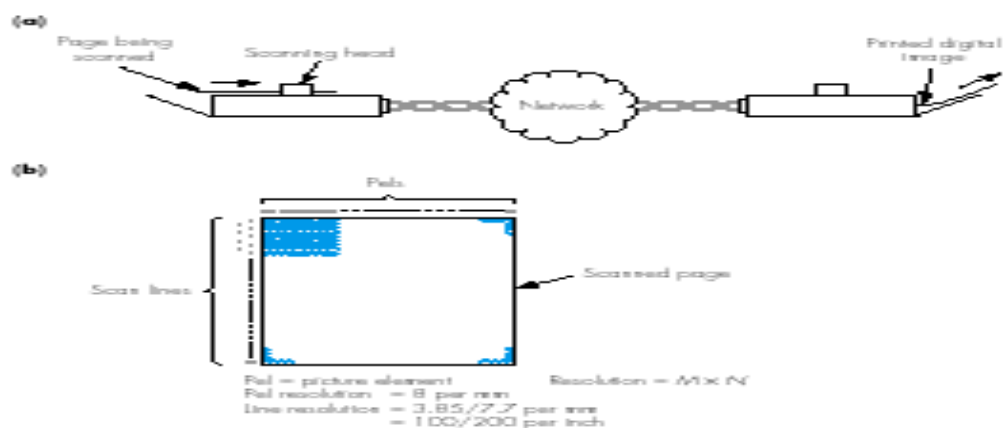
Figure 2.10 Graphics principles: (a) example screen format; (b) some simple object examples; (c) effect of changing position attribute; (d) solid objects.



Digitized documents

- Fig2.11
- A single binary digit to represent each pel, a 0 for a white pel and a 1 for a black pel

Figure 2.11 Facsimile machine principles: (a) schematic; (b) digitization format.



Digitized pictures

□ Color principles

□ A whole spectrum of colors—known as a color gamut—can be reproduced by using different proportions of red (R), green (G), and blue (B)

□ Fig 2.12

□ Additive color mixing producing a color image on a black surface

□ Subtractive color mixing for producing a color image on a white surface

□ Fig 2.13

Figure 2.12 Color derivation principles: (a) additive color mixing; (b) subtractive color mixing.

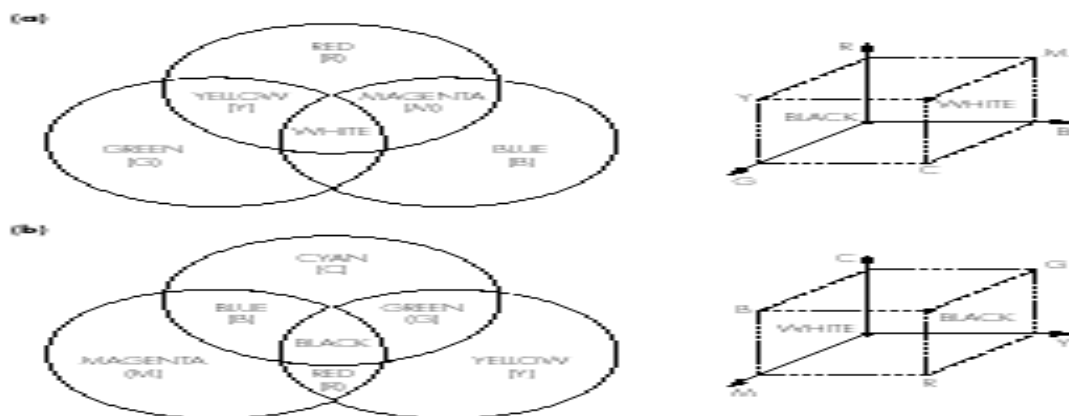
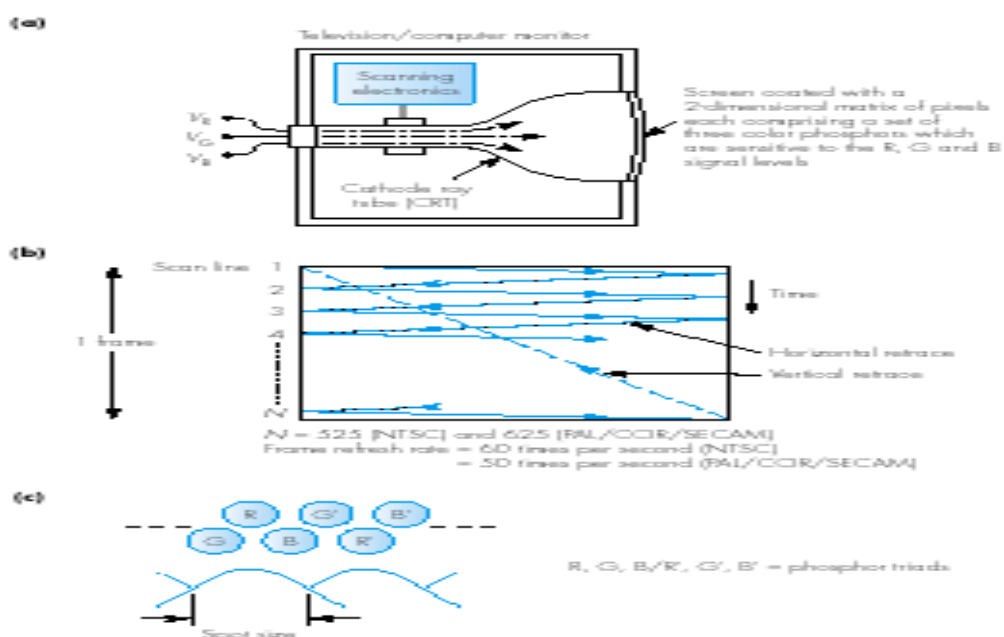


Figure 2.13 Television/computer monitor principles: (a) schematic; (b) raster-scan principles; (c) pixel format on each scan line.



2.4.3 Digitized pictures

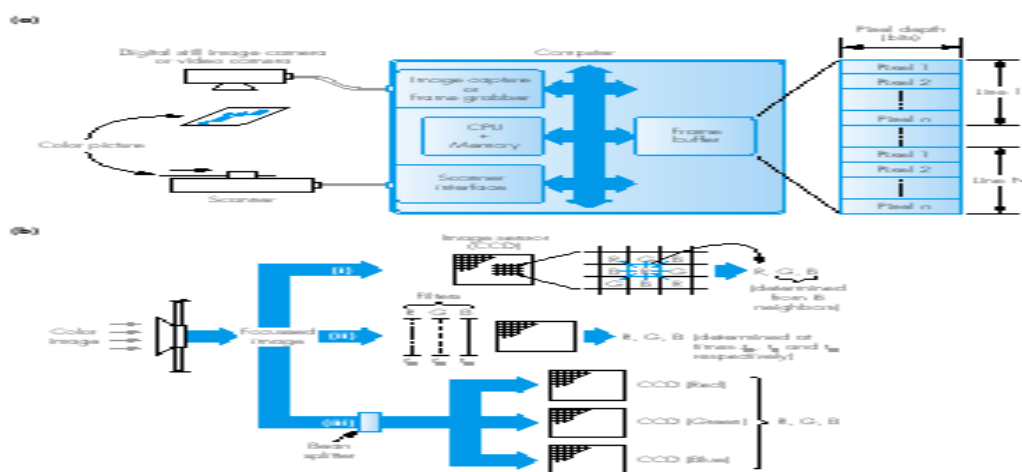
- Raster-scan principles
- Progressive scanning
- Each complete set of horizontal scan is called a frame
- The number of bits per pixel is known as the pixel depth and determines the range of different colors
- Aspect ratio
- Both the number of pixels per scanned line and the number of lines per frame
- The ratio of the screen width to the screen height
- National Television Standards Committee (NTSC), PAL(UK), CCIR(Germany), SECAM (France)
- Table 2.1

Table 2.1 Example display resolutions and memory requirements.

| Standard | Resolution | Number of colors | Memory required per frame (bytes) |
|----------|-----------------|------------------|-----------------------------------|
| VGA | 640 × 480 × 8 | 256 | 307.2 kB |
| XGA | 640 × 480 × 16 | 64K | 614.4 kB |
| | 1024 × 768 × 8 | 256 | 786.432 kB |
| SVGA | 800 × 600 × 16 | 64k | 960 kB |
| | 1024 × 768 × 8 | 256 | 786.432 kB |
| | 1024 × 768 × 24 | 16M | 2359.296 kB |

- Digital cameras and scanners
- An image is captured within the camera/scanner using an image sensor
- A two-dimensional grid of light-sensitive cells called photosites
- A widely-used image sensor is a charge-coupled device (CCD)
- Fig 2.16

Figure 2.16 Color image capture: (a) schematic; (b) RGB signal generation alternatives.



2.4.3 Digitized pictures

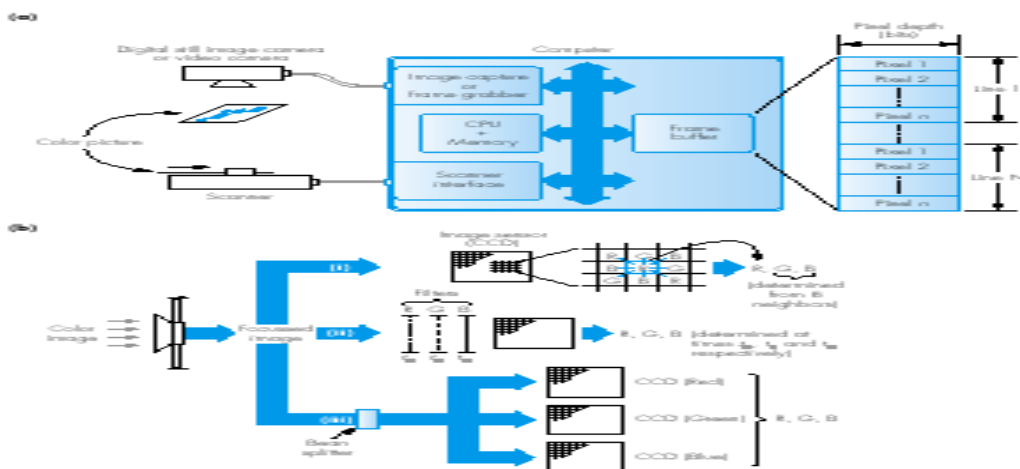
- Raster-scan principles
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Figure 2.16 Color image capture: (a) schematic; (b) RGB signal generation alternatives.



Audio

- The bandwidth of a typical speech signal is from 50 Hz through to 10 kHz; music

signal from 15 Hz through to 20kHz

- The sampling rate: 20ksps (2*10kHz) for speech and 40ksps (2*20kHz) for music
- Music stereophonic (stereo) results in a bit rate double that of a monaural (mono) signal
- Example 2.4

2.5.2 CD-quality audio

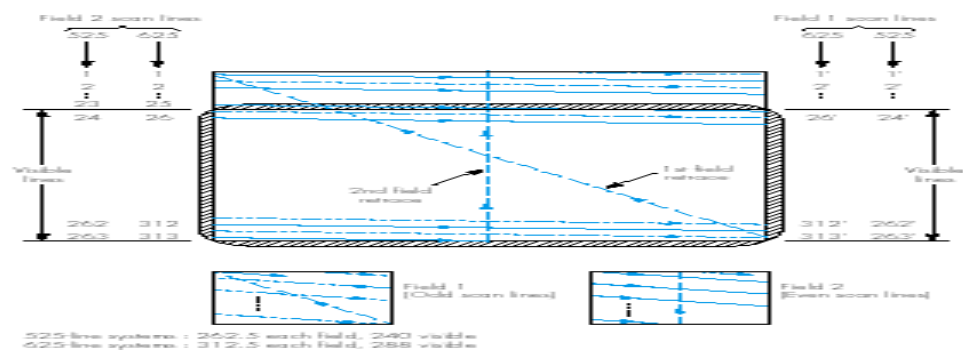
- Bitrate per channel = $\text{sampling rate} \times \text{bits per sample}$
 $= 44.1 \times 10^3 \times 16 = 705.6 \text{ kbps}$
- Total bitrate = $2 \times 705.6 = 1.411 \text{ Mbps}$
- Example 2.5

Video

Broadcast television

- Scanning sequence
- It is necessary to use a minimum refresh rate of 50 times per second to avoid flicker
- A refresh rate of 25 times per second is sufficient
- Field: the first comprising only the odd scan lines and the second the even scan lines
- The two fields are then integrated together in the television receiver using a technique known as interlaced scanning
- Fig 2.19
- The three main properties of a color source
 - Brightness
 - Hue: this represents the actual color of the source
 - Saturation: this represents the strength or vividness of the color

Figure 2.19 Interlaced scanning principles.



- The term luminance is used to refer to the brightness of a source

- The hue and saturation are referred to as its chrominance

$$Y_s = 0.299R_s + 0.587G_s + 0.144B_s$$

- Where Y_s is the amplitude of the luminance signal and R_s, G_s and B_s are the magnitudes of the three color component signals

- The blue chrominance (C_b), and the red chrominance (C_r) are then used to represent hue and saturation

- The two color difference signals:

$$C_b = B_s - Y_s$$

$$C_r = R_s - Y_s$$

- In the PAL system, C_b and C_r are referred to as U and V respectively

$$PAL: Y = 0.299R + 0.587G + 0.114B \quad U = 0.493(B - Y)$$

$$V = 0.877(R - Y)$$

- The NTSC system forms two different signals referred to as I and Q

$$NTSC: Y = 0.299R + 0.587G + 0.114B \quad I = 0.74(R - Y) - 0.27(B - Y)$$

$$Q = 0.48(R - Y) + 0.41(B - Y)$$

Digital video

- Eye has shown that the resolution of the eye is less sensitive for color than it is for luminance

- 4 : 2 : 2 format

- The original digitization format used in Recommendation CCIR-601

- A line sampling rate of 13.5 MHz for luminance and 6.75 MHz for the two chrominance signals

- The number of samples per line is increased to 720

- The corresponding number of samples for each of the two chrominance signals is 360 samples per active line

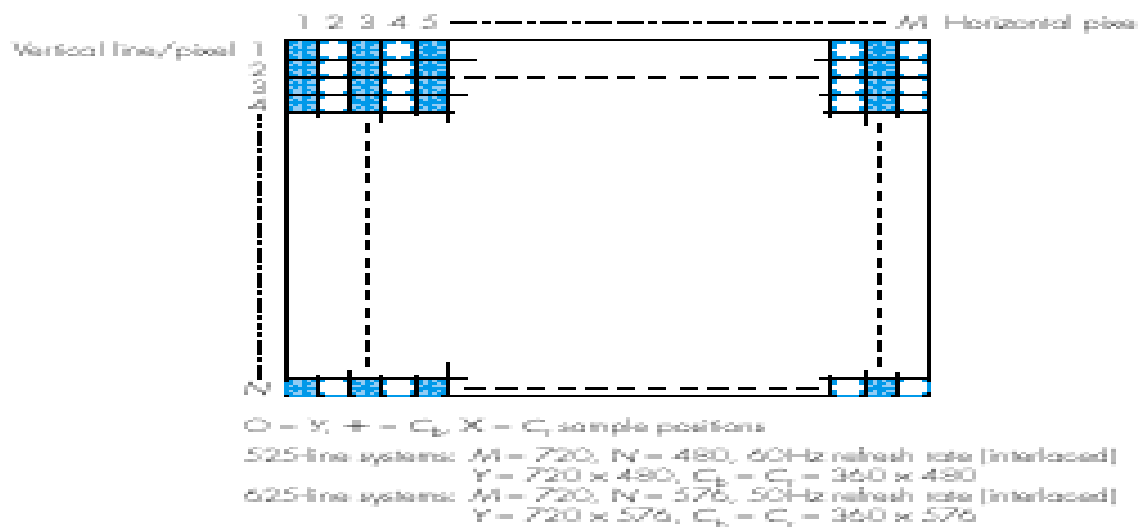
- This results in 4Y samples for every 2 C_b , and 2 C_r samples

- The numbers 480 and 576 being the number of active (visible) lines in the respective system

- Fig. 2.21

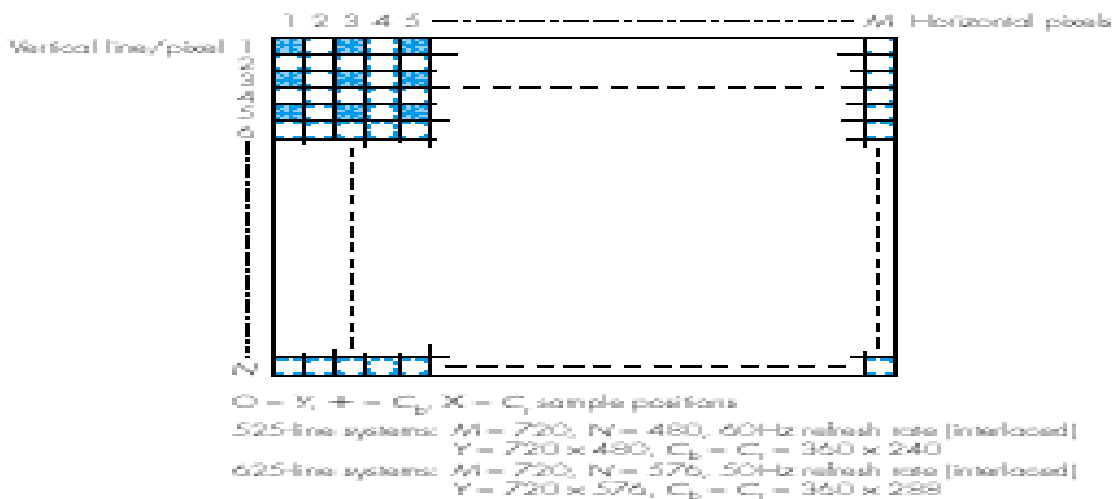
- Example 2.7

Figure2.21Samplepositionswith4:2:2digitizationformat.



- 4 : 2 : 0 format is used in digital video broadcast applications
- Interlaced scanning is used and the absence of chrominance samples in alternative lines
- The same luminance resolution but half the chrominance resolution
- Fig2.22

Figure2.22Samplepositionsin4:2:0digitizationformat.



525-line system

$$Y = 720 \times 480$$

$$C_b = C_r = 360 \times 240$$

625-line system

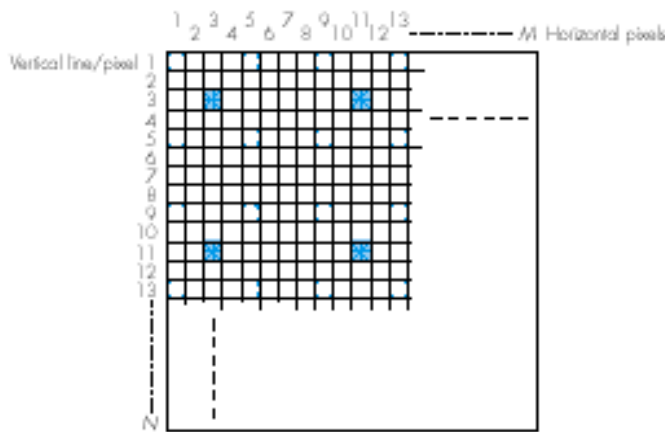
$$Y = 720 \times 480$$

$$C_b = C_r = 360 \times 288$$

$$13.5 \times 10^6 \times 8 + 2(3.375 \times 10^6 \times 8) = 162 \text{ Mbps}$$

- HDTVformats:theresolutiontothenewer16/9wide-screen tubes can be up to 1920*1152 pixels
- Thesourceintermediateformat(SIF)giveapicturequalitycomparablewith video recorders(VCRs)
- Thecommonintermediateformat(CIF)foruseinvideoconferencing applications
- Fig2.23
- ThequarterCIF(QCIF)foruseinvideotelephony applications
- **Fig2.24**
- **Table2.2**

Figure 2.23 Sample positions for SIF and CIF. Figure 2.24



O = Y, + = C_b , X = C_r sample positions
 $M = 720$, $N = 576$ with 15Hz or 7.5 Hz refresh rate (non-interlaced)
 $Y = 180 \times 144$, $C_b = C_r = 90 \times 72$

video digitization formats.

| System | Spatial resolution | Temporal resolution |
|----------|------------------------------|---------------------|
| 525-line | $Y = 640 \times 480$ | 60Hz |
| | $C_b = C_r = 320 \times 240$ | |
| 625-line | $Y = 768 \times 576$ | 50Hz |
| | $C_b = C_r = 384 \times 288$ | |
| 525-line | $Y = 320 \times 240$ | 30Hz |
| | $C_b = C_r = 160 \times 120$ | |
| 625-line | $Y = 384 \times 288$ | 25Hz |
| | $C_b = C_r = 192 \times 144$ | |
| | $Y = 384 \times 288$ | 30Hz |
| | $C_b = C_r = 192 \times 144$ | |
| | $Y = 192 \times 144$ | 15/7.5Hz |
| | $C_b = C_r = 96 \times 72$ | |

RECOMENDEDEDQUESTIONS:

1. Explain in code word, analog signal, signal encoder, signal decoder? [06]
2. Define—bandwidth & explain—band limiting channel? [05]
3. Explain Nyquist's sampling theorem & Nyquist rate? [04]
4. Define the meaning of term quantization interval & how it influences the accuracy of the sampling process of an analog signal? [06]
5. Explain
 - a) unformatted/plain text.
 - b) formatted/richtext.
 - c) Hypertext. [06]
6. Differentiate formatted text & unformatted text? Explain the acronym WYSIWYG? [05]
7. Explain briefly: visual object, freeform object, clipart, 3-D objects. [04]
8. Explain scanning, pels, digitization principles wrt to facsimile machines? [08]
9. Define the aspect ratio of a display screen. Give two examples of currently widely used screen sizes? [05]
10. Derive the time to transmit an image with each type of display assuming a bitrate of 56 kbps, 1.5 Mbps? [06]
11. Define text & image. [03]
12. Define audio & video. [03]
13. Compare formatted & unformatted text. [08]
14. What is rendering & clipart? [02]
15. What is flicker & frame refresh rate? [04]
16. What is NTSC & PAL? [10]
17. What is sample & hold, Quantizer? [06]
18. Define aspect ratio & pixel depth. [06]